

IN THE SPECIFICATION:

Please replace the last paragraph of page 7, continued on page 8, with the following replacement paragraph:

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AI  
Figure 3A illustrates the initial steps that are performed by user interface resources 14a and 14b according to the first and second embodiments, respectively. Typically a subscriber accesses his or her user interface session by placing a call to the user interface resource 14 in step 30 40. In particular, assume that the subscriber 20a places a voice over IP call to the user interface resource 14 via the IP telephony gateway 12; the user interface resource 14 establishes an RTP data stream 16a for a user interface session with the subscriber 20a by providing signaling instructions to the IP telephony gateway 12. The IP telephony gateway 12 in response supplies the RTP data stream 16a to the user interface resource 14, establishing a user interface session between the user interface resource and the subscriber 20a for intelligent dial tone services such as centralized management of unified messages (e.g., voice, fax, or e-mail messages), voice-activated retrieval of web content, or voice activated dialing.

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Please replace the last paragraph of page 8, continued on page 9, with the following replacement paragraph:

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A2  
The user interface resource 14, in response to receiving a call request from the subscriber 20a via the RTP stream 16a in step 32, utilizes speech recognition resources in step 34 to interpret the call request. The user interface resource 14, in response to determining that the request is for initiating an outgoing call, interprets any spoken number within the call request to identify the destination number in step 36, or alternately obtains in step 38 the destination number from the list of stored numbers in the subscriber database 24 based on the spoken name interpret by the speech recognition resource. The user interface

resource 14 then initiates in step ~~40~~ 42 a second RTP data stream 16b to the destination party 20b based on the destination number obtained from steps 36 or 38. As part of the call setup and establishment of the second RTP stream 16b, the user interface resource 14 communicates with the H.323 resource within the IP telephony gateway 12 and stores the H.245 protocol capabilities (e.g., what codec to use, port number for sending media streams, etc.) from the IP telephony gateway 12 for the outbound call 22b in step 44. The IP telephony gateway 12 then establishes a connection via the call leg 22b and the second RTP stream 16b between the destination party 20b and the user interface resource 14, enabling the user interface resource 14 to determine whether the first RTP stream 16a and the second RTP stream 16b should be bridged.

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